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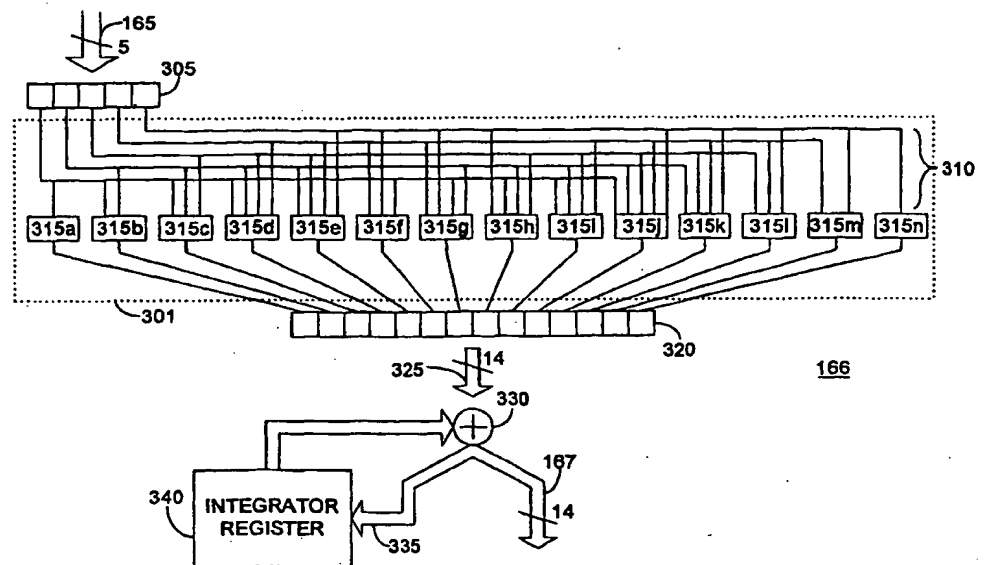
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(21) International Application Number: <b>PCT/US97/21469</b> (22) International Filing Date: <b>21 November 1997 (21.11.97)</b> (30) Priority Data: 60/031,350           21 November 1996 (21.11.96)   US Not furnished       12 November 1997 (12.11.97)   US (71) Applicant: <b>ADVANCED MICRO DEVICES, INC. [US/US];</b> 5204 East Ben White Boulevard, Mail Stop 562, Austin, TX 78741 (US). (72) Inventors: <b>JANESCH, Stephen, T.; 4106 Reynosa Drive,</b> Austin, TX 78739 (US). <b>SCHINZLEIN, Paul; 11410 Antler</b> Bend Road, Austin, TX 78737 (US). (74) Agent: <b>RILEY, Loius, A.; Advanced Micro Devices, Inc.,</b> 5204 East Ben White Boulevard, Mail Stop 562, Austin, TX 78741 (US).		(81) Designated States: JP, KR, European patent (AT, BE, CH, DE, DK, ES, FI, FR, GB, GR, IE, IT, LU, MC, NL, PT, SE).  <b>Published</b> <i>With international search report.</i> <i>Before the expiration of the time limit for amending the claims and to be republished in the event of the receipt of amendments.</i>

(54) Title: A PROGRAMMABLE LOOP FILTER FOR CARRIER RECOVERY IN A RADIO RECEIVER

## (57) Abstract

A digital loop filter in the carrier-recovery loop of a digital communications receiver. The recovery loop is a PLL that keeps the receiver oscillator locked to the carrier wave, and the loop filter provides control over the PLL's frequency response by conditioning an error signal that is fed back to the receiver oscillator. In the present invention, the error signal is a digital signal, and the loop filter is implemented in digital hardware. With this implementation the characteristics of the loop filter are determined by logic design rather than by physical features of analog components, thereby giving this filter a more precise function than one with analog integrators.

This implementation is also immune to the low tolerances typical of the manufacturing process for analog devices (especially on monolithic circuits), and is more easily adjusted than its analog counterparts. Two gain coefficients characterize the loop filter in the present invention. These gain coefficients are chosen to be powers of two, simplifying the process of multiplying them with the digital error signal. The gain coefficients are read from a memory, making the loop filter easily programmable. By changing the gain coefficients during operation of the receiver, the carrier-recovery loop can be placed in one of the several operating modes, including acquisition, tracking, and hold. The receiver can be configured with the appropriate values of the gain coefficients for each operating mode during the initial assembly and during subsequent reconfigurations.



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**TITLE: A PROGRAMMABLE LOOP FILTER FOR CARRIER RECOVERY IN A RADIO RECEIVER**

**BACKGROUND OF THE INVENTION**

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Field of the Invention

The invention relates to electronic communication and, more particularly, to a loop filter in the carrier-recovery loop of a radio receiver.

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Description of the Related Art

Radio transmission and reception is accomplished through a carrier wave that is modulated to bear the transmitted information. The transmission of the data involves modulating the carrier with a baseband signal that represents the information to be transmitted. Typically, the carrier wave is generated by a reference oscillator in a transmitter unit and modulated by a modulator to produce the transmitted signal. After traversing a communication channel, this signal is received by a receiver unit that demodulates it to extract the baseband signal.

An important component of the receiver unit is a local oscillator that is used to demodulate the received signal. This oscillator must match the frequency of the transmitter oscillator that generated the carrier wave: if the frequencies of the two oscillators are not matched, the receiver cannot efficiently demodulate the transmitted signal. The receiver oscillator can be built so that its natural frequency is close to that of the transmitter oscillator, but due to variations in manufacturing and differences in operating environments there will be drifts between the two oscillators. To compensate for such offsets in frequency between the carrier wave and the receiver oscillator, the receiver oscillator can be locked to the carrier wave by incorporating it into a phase-locked loop (PLL). Such a PLL serves as a carrier-recovery loop that ties the frequency of the receiver oscillator to the frequency of the transmitter oscillator.

In addition to the receiver oscillator, the carrier-recovery loop includes a phase detector and a loop filter. The phase detector generates an error signal to represent the difference in phase between the receiver oscillator and the carrier wave. Since the original carrier wave is not typically available to the receiver unit, the phase detector must be able to extract the frequency of the carrier wave from the received signal. That is, it must be able to ignore variations in the received signal's phase that are due to the information encoded onto the carrier. For example, in the case of a digital communication system using differential-quadrature-phase-shift-keying (DQPSK) modulation, changes in the phase of the carrier by multiples of 90° must not be interpreted as a drift in the receiver oscillator's phase. Depending on the type of modulation, there are several established methods of making the phase detector in the carrier-recovery loop insensitive to the phase shifts due to data-bearing modulation.

The loop filter in the carrier-recovery loop receives the error signal from the phase detector. The error signal is filtered in the loop filter into a feedback signal. The feedback signal is then used to adjust the frequency of the receiver oscillator so that it tracks the frequency of the received signal. The filtering typically includes a low-pass filtering characterized by several gain coefficients that determine the speed and sensitivity

of the PLL. Large gain coefficients lead to a fast PLL, which reduces the time lag for the receiver oscillator to track the carrier wave. However, with the faster PLL comes a reduced robustness of the lock: a faster PLL is more susceptible to having its oscillator's phase shifted out of lock by noise in the received signal. Once the lock is lost it can be reacquired, but it may have a phase error (of  $2n\pi$ ) called a cycle-slip.

Since a fast PLL and a low incidence of cycle-slips are both desirable qualities, the desirable values for the gain coefficients are trade-offs between speed and robustness. There are several factors that determine the desired values of the gain coefficients. In qualitative terms, high gain coefficients (leading to a fast PLL) are appropriate if the received signal has a stable, slowly varying frequency that is close to the frequency of the receiver oscillator. This is the case for low-noise transmissions when the receiver oscillator is already locked to a good received signal. For these signals, a fast PLL keeps the receiver oscillator tightly locked to the received signal. There is, however, an upper limit on the speed of the PLL because the faster its response, the more susceptible it is to cycle slips. In general, lower PLL speeds are required for noisier received signals.

The gain coefficients also need to be adjusted as the recovery loop switches between different operating modes. The previously described tradeoff between PLL speed and sensitivity to noise applies when the PLL is tracking the received signal. There is an appropriate range of values for the gain coefficients in this *tracking mode*. Another mode of operation for the PLL is when it initially acquires a phase lock to a received signal. During this *acquisition mode*, higher values of the gain coefficients are necessary so that the receiver oscillator can quickly approach the frequency of the received signal. A third mode of operation is the *hold mode*, when the receiver oscillator is kept at a fixed frequency, ignoring the received signal. This mode is desirable, for example during temporary losses of the received signal during fades. In the hold mode, some or all of the gain coefficients are zeroed so that no new feedback is provided to the receiver oscillator.

The determination of the quantitative values for the gain coefficients depends on the amplitudes of the received signal, the noise in the received signal, and the hardware used in the implementation of the PLL. None of these factors can be perfectly pre-determined. In addition it depends on the operating mode of the recovery loop. The received amplitudes and noise will vary with the conditions of the receiver's use, the hardware is subject to variations in manufacturing processes, and the recovery loop will switch between different modes during operation. Because of these variations, PLLs are generally made with an array of loop filters with different combinations of gain coefficients. The filter with the most appropriate gain coefficients is selected during operation of the receiver to place the recovery loop in an appropriate operating mode.

A loop filter typically generates an output that is a linear combination of two components: the input phase error signal and the time-integral of this input. Therefore, some form of an integrator is a standard component of the loop filter, and the integrator's time constant determines one of the filter's gain coefficients. The integrators are typically analog devices that rely on the physical properties and dimensions of their components to determine their outputs. These parameters can vary under different operating conditions, making the output a less controllable signal and introducing a limitation on the prior art carrier-recovery loops. The limited selection and low tolerance of these analog components in a filter circuit limit the flexibility and tolerance of the gain coefficients. Therefore, an improved loop filter is desired for a carrier-recovery loop with adjustable and well-defined gain coefficients.

### **SUMMARY OF THE INVENTION**

The present invention comprises a loop filter for a carrier-recovery loop in the receiver of a communication system. In the present invention, the error signal generated by the phase detector and the feedback signal from the loop filter are both digital signals. Since both its input and output are digital, the loop filter of the present invention is implemented in digital hardware, which brings several advantages to the design.

Unlike integration performed by analog components on integrated circuits, digital integration is a precisely controllable function determined by logic design rather than the physical features of its components. Analog implementations of integrators suffer from low tolerances in the manufacturing process; this is especially true in monolithic integrated circuits. An analog integrator in one circuit may have a very different time constant than an integrator in another circuit manufactured by the same process. In contrast, all digital integrators produced by the same process will have essentially the same performance characteristics.

Another advantage of the invention is that digital integrators are more flexible than their analog counterparts. While an analog integrator requires a selection of reference resistors and/or capacitors to provide multiple time constants, a digital integrator can be easily programmed to change its function.

Two gain coefficients, designated  $k_1$  and  $k_2$ , characterize the loop filter in the present invention. These gain coefficients determine the frequency response of the loop filter and the amplitude of its output. Explicitly, the output feedback signal is a sum of two components: the input error signal, multiplied by  $k_1$ , and the integral of the input error signal, multiplied by  $k_2$ . The two gain coefficients are chosen to be powers of two, simplifying the process of multiplying the digital error signal by them.

The loop filter reads gain coefficients  $k_1$  and  $k_2$  from a memory where they are stored; this arrangement makes the loop filter easily programmable since the values of the gain coefficients can be readily changed in the memory. By changing the gain coefficients during operation of the receiver, the recovery loop can be selectively placed in one of the several operating modes, including tracking, acquisition, and hold. The receiver can be configured with the appropriate values of the gain coefficients for each operating mode during the initial assembly and during subsequent reconfigurations.

While being simple in design and implementation, this system is a flexible filter that is adaptable to a range of receiver designs and constructions, and tolerant of a range of amplitudes for the received signal. The present invention thus provides a significant improvement and advance in the art and technology of carrier-recovery loop filters.

### **BRIEF DESCRIPTION OF THE DRAWINGS**

Other objects and advantages of the invention will become apparent upon reading the following detailed description and upon reference to the accompanying drawings in which:

Fig. 1 is a schematic view of a telecommunication system;

Fig. 2 shows the carrier-recovery loop of the receiver in Fig. 1;

Fig. 3 is a block diagram of one embodiment of the loop filter in the carrier-recovery loop of Fig. 2;

Fig. 4 is a block diagram of a selection unit in the loop filter of Fig. 3;

Fig. 5 is a flowchart for the carrier-recovery loop of Fig. 2; and

Fig. 6 is a flowchart describing the loop filter of Fig. 3.

### **DESCRIPTION OF THE PREFERRED EMBODIMENTS**

#### **Digital Communication System**

As shown in Fig. 1, a digital communication system comprises at least one transmitter 100 and one receiver 150 for the communication of data. Such communication systems are well known in the art. The system described in this figure uses differential quadrature phase-shift keying (DQPSK) to convey data from a transmitter to a receiver. Although not depicted here, other modulation schemes such as ASK, FSK, and other variants of PSK could also be used to convey the data.

In the transmitter 100, digital data 102 are provided to a modulator 106. A transmitter reference oscillator 104 generates a sinusoidal carrier wave 105 for the modulator 106. The digital data 102 are encoded onto the sinusoidal carrier wave 105 by the modulator 106 which shifts the carrier's phase by multiples of  $90^\circ$  according to the technique of DQPSK modulation, a technique well-known in the art. In this technique, the modulator 106 shifts the phase of the carrier wave by multiples of  $90^\circ$  to generate a transmitted signal 108; these phase shifts are the symbols that encode the data. Each symbol lasts for a duration of time  $T$  after which the next phase shift is introduced to the carrier. The differences in phase angle between successive symbols represent the transmitted data 102. Since there are four possible symbols (shifts of  $0^\circ$ ,  $90^\circ$ ,  $180^\circ$ , or  $270^\circ$ ) in QPSK modulation, each phase difference represents two bits of the transmitted data.

The carrier wave's frequency is determined by the reference oscillator 104 in the transmitter. The transmitted signal 108 is the sinusoidal carrier wave with the data-bearing phase shifts of duration  $T$ . The transmitted signal 108 is sent via a physical communication channel 190 to the receiver 150.

The channel depicted in this figure is a radio transmission system that modulates the transmitted signal onto a radio wave 194 with a frequency greater than the carrier wave frequency. The channel 190 depicted here comprises the radio-frequency (RF) modulator 192, the radio wave 194 transmitted through the air, and the RF demodulator 196. As would be known to one skilled in the art, other communications channels such as transmission line, waveguide, or optical fiber systems can of course be used instead of (or in conjunction with) the depicted radio transmission system. Under ideal conditions the received signal 158 would be an exact replica of the transmitted signal 108. In practice, however, there may be some differences between these two signals due to degradation suffered in the communication channel.

In the receiver 150, the received signal is demodulated by a demodulator 156 to extract the received data 152. Ideally, the received digital data 152 would replicate the transmitted digital data 102, but in practice the two sets of data may differ due to decoding errors in the receiver, or degradation of the transmitted signal in the communications channel.

### Carrier Recovery

To extract the data from the received signal, the demodulator 156 requires a reference signal that closely reproduces the carrier wave 105. Since the original carrier wave 105 is not usually available in the receiver unit, this reference 155 is generated by a reference oscillator 154 in the receiver. In a preferred embodiment of the receiver, the reference oscillator 154 is a digitally controlled oscillator (DCO); that is, it accepts a digital input word that controls the frequency of the oscillator's output. This oscillator 154 must match the frequency of the transmitter oscillator 104 that generated the carrier wave 105; if the frequencies of the two oscillators are not matched, the receiver unit 150 cannot efficiently demodulate the transmitted signal. The receiver oscillator 154 can be built so that its natural frequency is close to that of the transmitter oscillator 104, but due to variations in manufacturing and differences in operating environments there will be drifts between the two oscillators. To compensate for such offsets in frequency between the carrier wave and the receiver oscillator, the receiver oscillator is locked to the carrier wave by incorporating it into phase-locked loop (PLL). The PLL is a carrier-recovery loop 162 that ties the frequency of the receiver oscillator 154 to the frequency of the transmitter oscillator 104. The feedback from the carrier-recovery loop 162 corrects offsets between the frequencies of the receiver oscillator and the carrier.

The depiction of the receiver in Fig. 1 includes a basic block diagram of the carrier-recovery loop 162. The carrier-recovery loop 162 includes the basic elements of a PLL: the receiver oscillator 154, a phase detector 164, and the loop filter 166.

Fig. 2 shows an embodiment of the carrier-recovery loop 162. The phase detector 164 receives the received signal 158 and the receiver reference signal 155. With these two inputs, the phase detector 164 compares the receiver oscillator's phase to the phase of the carrier wave and generates a digital phase error signal 165 indicative of the phase shift between them. The phase error signal 165 is then provided to the loop filter 166 which comprises a novel configuration as described below. The loop filter 166 uses digital processing elements to condition the phase error signal 165 to generate a feedback signal 167; this feedback signal is fed back to the digitally controlled receiver oscillator 154 to nullify its offset from the carrier frequency.

In the implementation of the carrier-recovery loop presented in this figure, the digital feedback signal is fed back to the receiver oscillator 154, which produces the receiver reference signal 155. The receiver reference signal 155 is made available to the phase detector for comparison with the received signal 158.

### The Loop Filter

As shown in Fig. 2, the loop filter 166 comprises a multiplier 201 with a gain coefficient  $k_1$ , which receives the phase error signal 165 and provides an output to a digital adder 204. The loop filter 166 also comprises a multiplier 202 with a gain coefficient  $k_2$ , which receives the phase error signal 165 and provides

an output to an integrator 203 in the loop filter. The integrator 203 in turn provides an output to the digital adder 204. The digital adder 204 provides the sum of its two inputs to the receiver oscillator 154.

The gain coefficients  $k_1$  and  $k_2$  in multipliers 201 and 202 are adjustable binary values stored in a memory 210.

The integrator 203 accumulates the value of the phase error signal 165 after it has been scaled by the gain coefficient  $k_2$  in multiplier 202. The digital adder 204 combines this integrated signal with a version of the original phase error signal that has been scaled by the gain coefficient  $k_1$  in multiplier 201. Thus the complex-frequency transfer function of the loop filter is  $k_1 + k_2/s$ . With this implementation of the loop filter, the complex-frequency transfer function  $H_\theta(s)$  for the full carrier-recovery loop is given by the following equation.

$$H_\theta(s) = \frac{\Phi(s)}{\Theta(s)} = \frac{sk_1 + k_2}{s^2 + sk_1 + k_2} \quad \text{Eq. 1}$$

Here  $\Phi(s)$  represents the phase of the receiver oscillator (in the complex-frequency domain), and  $\Theta(s)$  represents the phase of the received signal 158. The PLL thus has a low-pass response to changes in input frequency. The time constant for its response is determined by the gain coefficients  $k_1$  and  $k_2$ . Since the gain coefficients are binary values stored in the memory 210, they can be adjusted to put the carrier-recovery loop into one of several different operating modes. In the present invention, the receiver has three operating modes: acquisition, tracking, and hold.

To enter the acquisition mode, the receiver sets these coefficients to the appropriate acquisition values each time the receiver begins carrier recovery. In acquisition mode, the PLL of the preferred embodiment has a low-pass response to input frequency change. The receiver changes from the acquisition to tracking mode by reprogramming the gain coefficients  $k_1$  and  $k_2$  in the loop filter to lower values that are appropriate when the oscillator is close in frequency to the received signal. In tracking mode, the values of  $k_1$  and  $k_2$  are reduced so that the PLL slows its response time, thereby reducing its sensitivity to high-frequency noise.

This change from acquisition mode to tracking mode occurs when the receiver oscillator is determined to be adequately matched to the frequency of the received signal. There are several possible criteria for changing between these modes. One criterion for making this switch from acquisition mode to tracking mode is that the recovered frequency should be within a set range (typically 1 kHz) of the actual input frequency. A second requirement is useful in systems that receive digital data. In these systems, the switch to tracking mode can be additionally delayed until the receiver has acquired a frame synchronization with the received signal.

In hold mode the receiver oscillator is not allowed to adapt, so that it continues to produce its last known frequency. This mode is used to sustain the appropriate frequency during fades in the received signal. In this mode the gain coefficients  $k_1$  and  $k_2$  have values of zero. Alternatively, this mode can be accomplished by holding the value of the digital feedback signal constant or by forcing the (phase error) input to the loop filter to zero. The latter means can be used to conserve power in TDD (time-division duplex) communication



systems. It allows the clock to the multipliers for  $k_1$  and  $k_2$  and the integrator to be stopped during the transmit portion of the TDD frame, reducing their power requirements by up to a factor of two.

The block diagram in Fig. 3 shows an implementation of the preferred embodiment of the loop filter 166. In this implementation, an input register 305 receives a 5-bit number representing the digital phase error signal 165. The bits of this number are sent to a multiplier 301 that multiplies them by the gain coefficients  $k_1$  and  $k_2$ . Since these gain coefficients are powers of 2, the multiplier works by shifting the input by an appropriate number of bits, as described below, to generate a 14-bit product 325 in an output register 320. The multiplier alternates between using  $k_1$  and  $k_2$  to multiply the phase error signal 305, so the product 325 represents the phase error multiplied by  $k_1$  on one clock cycle, and then the phase error multiplied by  $k_2$  on the next clock cycle.

The product 325 is sent to a time-multiplexed adder 330 that alternates its function from cycle to cycle. During a cycle in which it receives the product of  $k_1$  and the phase error signal from the multiplier 301, it adds this product to a value that it receives from an integrator register 340. The resulting sum is a 14-bit number representing the feedback signal 167, which is the output of the loop filter. On alternate cycles, the time-multiplexed adder 330 receives the product of  $k_2$  and the phase error signal from the multiplier 301. During these cycles it adds the product to the value it receives from an integrator register 340; this sum 335 is then sent back to the integrator register 340 and is stored there. This implementation realizes the function of the loop filter 166 that was described in the discussion of Fig. 2.

When the carrier recovery loop is in the hold mode, the input register is simply set to contain all zeroes, regardless of the value of the phase error 165. The resulting feedback signal 167 is then also zero, as required for the hold mode. For the other two modes, tracking and acquisition, the multiplier multiplies the phase error signal 165 by the appropriate values of  $k_1$  and  $k_2$ .

The multiplier 301 is implemented by the connections and elements contained in the dashed box in Fig. 3. The five bits in the input register 305 are sent via a set of connections 310 to fourteen selection units 315a-n. These selection units are each coupled to one of the bits in the multiplier's output register, and they each copy either one of the bits from the input register or a zero into their corresponding bit in the output register. The different selection units are configured so that the output register receives a copy of the bits in the input register, but shifted by the appropriate number of places according to the multiplier ( $k_1$  or  $k_2$ ).

Fig. 4 presents a more detailed block diagram of the selection units 315a-n in the multiplier 301. As described earlier, each selection unit receives one or more bits from the input register 305 via the a set of connections 310. In this figure, these bits are shown as the binary inputs 410 for one of the selection units. The inputs are sent to a multiplexer 415 which selects one of them, or a zero 411, as the selection unit's output 420. This output is sent to one of the bits in the multiplier's output register (as was shown in Fig. 3).

A logic block 440 controls the multiplexers 415 in the selection units 315a-n by generating a shift code 460 that determines which of the binary inputs 410 is selected by each multiplexer 415. The logic block 440 chooses which input bit is selected by the multiplexer so that the multiplier output register 320 receives an appropriately shifted copy of the input register 305.

To determine the number of places by which the multiplier input 165 should be shifted, the logic block receives the four different values of the multiplier: the value of  $k_1$  for lock mode 431, the value of  $k_1$  for

acquisition mode 432, the value of  $k_2$  for lock mode 433, and the value of  $k_2$  for acquisition mode 434. These values are pre-programmed into a memory as appropriate for the different modes. Another input 450 to the logic block 440 indicates the cycle in the time-multiplexing; that is, whether  $k_1$  or  $k_2$  is being used as a multiplier. The logic block 440 also has an input 455 that indicates the operating mode of the carrier recovery loop: lock or acquisition. These two inputs 450 and 455 determine which of the four multiplier values 431-434 is used by the logic block 440. With this multiplier value, the logic block 440 generates the shift code 460 and provides it to the multiplexer 415. In response to the shift code 460, the multiplexer 415 selects one of the input bits 410 from the input register 305 or a zero 411 as the selection unit output 420 that is sent to the corresponding bit of the multiplier output register 320.

In this preferred embodiment, the output of the logic block 440 indicates the shift code, that is, the number of bits to shift the input value 165 stored in register 305. The logic needed in block 440 to generate the shift code is easily implemented by one skilled in the art of logic design, and the multiplexer arrangement is well known as a "shifter" or "barrel shifter". The constraint that each of  $k_1$  and  $k_2$  be a power of 2 allows the shifter to function as a multiplier. This embodiment of the invention utilizes factors  $k_1$  and  $k_2$  that are less than 1, thus negative powers of 2 ( $k_1, k_2 = 2^n$ ;  $n = -1, -2, -3, \dots$ ). However, in another embodiment of the present invention, the factors  $k_1$  and  $k_2$  can also take values that are greater than or equal to 1 ( $k_1, k_2 = 2^n$ ;  $n = 0, \pm 1, \pm 2, \pm 3, \dots$ ), and the shift codes are chosen appropriately.

The sequence of steps which constitute the carrier recovery are illustrated by the flowchart in Fig. 5. In this diagram, the bold blocks and flow-lines on the right indicate the steps and the flow between them, while the light boxes on the left indicate quantities that are calculated and used in these steps. From the starting conditions 501, the first step 503 is to program the appropriate gain coefficients for the current operating mode into the memory 210. The next step is to receive the received signal 505. The error signal 165 representing the offset of the receiver oscillator 154 (shown in Fig. 1 and Fig. 2) is generated in the next step 510. The error signal 165 is filtered in the following step 515 according to the gain coefficients  $k_1$  and  $k_2$  received from memory 530. This filtering step 515 generates the feedback signal 167 that is used in the next step 520 to adjust the receiver oscillator so that it better matches the received signal. After this adjustment, the recovery loop returns to its initial step 505 to repeat the procedure with a new sample of the received signal.

The step 515 of filtering the error signal is expanded in Fig. 6. Here the bold blocks and flow-lines in the middle of this figure indicate the filtering steps and the flow between them, while the light boxes on the left and right indicate quantities that are calculated and used in these steps. The start of the filtering procedure 601 is right after the error signal has been generated 510 (as was shown in Fig. 5). In the first step 610 of the filtering procedure the error signal 165 is multiplied by the gain coefficient  $k_1$  received from memory 530 to generate the product 650 of these two quantities. The product 650 is then added in the next step 615 with a signal 660 representing the integral of the error signal 165 multiplied by the gain coefficient  $k_2$ . The resulting sum is the digital feedback signal 167, which is the output of the filtering procedure 515.

To update the integrated signal 660, the next step 620 in the filtering procedure 515 multiplies the error signal 165 with the gain coefficient  $k_2$  received from memory 530. The resulting product 655 is then added 625 to the integrated signal 660. The resulting sum, which represents the incremented value of the integrated signal 660, replaces the old value of the integrated signal 660.

Having thus generated an updated value for the feedback signal 167 and the integrated signal 660, the filtering procedure comes to a termination 699, and the carrier recovery of Fig. 5 proceeds to adjust the receiver oscillator in step 520.

#### 5 Digital Compensation of TDD Frequency Pulling

In one embodiment of the invention, the integrated signal 660 is stored in a memory 665 (shown in Fig. 6) which then holds a stored integrator value. The value in this memory 665 is then used to as an initializing value for the integrated signal 660. The memory 665 is operable to provide its value back to the integrated signal 660 when the carrier recovery loop begins to acquire a new phase lock. This feature of storing the integrated signal 660 in a memory 665 is especially useful in carrier recovery loops that are incorporated into some time-division duplexing (TDD) or time-division multiple access (TDMA) transceivers, in which a unit alternates between receiving and transmitting data. It is a well-known problem in TDD and TDMA radio architecture to have frequency shifts in reference oscillators between transmission and reception modes. This frequency pulling occurs due to operating differences between the transmission and reception modes, such as changes in the output impedance of a reference oscillator.

In another embodiment of the invention, the receiver 150 and the RF demodulator 196 from Fig. 1 are incorporated in a TDD radio transceiver along with a local transmitter and a local RF modulator. In this embodiment, an RF oscillator in the RF demodulator 196 is used to demodulate the RF signal 194 during reception, and is also used by the local RF modulator to generate an RF carrier during transmission. The frequency of this oscillator undergoes transient frequency shifts as shown in Fig. 7a. In this graph, the RF oscillator frequency is plotted versus time over the duration of a TDD frame. The vertical axis on the left of the figure indicates a center RF value  $f_0$ . In this embodiment, the oscillator frequency, shown by the light curve, has a rapid positive jump when the transceiver begins to receive data. The RF oscillator frequency gradually returns to  $f_0$ , then suffers a rapid negative jump as the transceiver switches to transmit data, and again gradually returns to  $f_0$ . This pulling of the RF oscillator frequency can lead to significant data losses if it is not compensated, since the large frequency shifts place significant demands on the carrier-recovery loop 162. If the carrier recovery loop fails to track the frequency shifts for a portion of the data reception, the received data will be lost for that portion of the reception. Traditionally, this pulling has been compensated by RF design modifications of the RF demodulators 162 in the prior art. However, in this embodiment of the invention, the stored integrator value in the memory 665 can be used to remedy the effects of the RF frequency pulling.

The heavy curve in Fig. 7a illustrates an example of the recovered frequency for the DCO 154 in this TDD embodiment of the invention. The recovered frequency is at an intermediate frequency (IF) that is lower than the frequency of the RF signal 194, but variations in the RF oscillator frequency lead to corresponding variations in the recovered frequency. The vertical axis for the recovered frequency is on the right in the figure. The center value  $f_0$  on this axis indicates the corresponding phase-locked recovered frequency when the RF oscillator is at  $f_0$ . Variations in the RF oscillator lead to a corresponding hertz-for-hertz variation in the locked recovered frequency. Thus, on this graph, the heavy curve showing the recovered frequency would lie on top of the light curve showing the RF oscillator frequency if the carrier-recovery loop 162 were ideally tracking the frequency-pulling of the RF oscillator. This would not, however, be the optimal condition for carrier recovery,

since if the loop 162 is fast enough to track such a large sudden shift, then it may be too susceptible to high-frequency noise to maintain an adequate lock.

Instead, this TDD embodiment of the invention uses the digital word stored in the memory 665 to mitigate the effects of TDD frequency pulling. As shown by the heavy curve in Fig. 7a, over the duration of a TDD frame, the recovered frequency starts at some initial value  $f_1'$  and may have a significant offset from the received signal as the carrier-recovery loop attempts to match the sudden change in received frequency. The TDD frames are structured so that no payload data are transmitted during the initial portion of each transmitted data frame. This portion, called the preamble, is used to allow feedback loops to settle during the initial reception of a data frame. At the end of the preamble, the recovered signal has come closer to its target value (of lying on top of the RF oscillator curve). The exact curve followed by the recovered frequency depends on the form of the frequency pulling and on the response characteristics of the carrier recovery loop 162. The recovered frequency  $f_2'$  at the end of the preamble is stored in memory 665 and is used as the initializing value for the recovery frequency at the beginning of the next received frame. As shown in Fig. 7b, by starting from this improved initial value  $f_2'$ , the carrier-recovery loop more quickly approaches a phase lock during the second received frame. At the end of the preamble in the second received frame, the memory 665 stores a further-improved approximation  $f_3'$  of the starting recovered frequency. This new approximation is used in the following frame, as shown in Fig. 7c. Thus by building on values previously stored in memory 665, the carrier-recovery loop 162 converges on a good initializing value for the start of a received data frame. This initializing value allows the carrier-recovery loop 162 to maintain a tight lock with the received signal 158 despite the transient frequency jumps in the RF oscillator of the RF demodulator 196.

It is to be understood that multiple variations, changes and modifications are possible in the aforementioned embodiments of the invention described herein. Although certain illustrative embodiments of the invention have been shown and described here, a wide range of modification, change, and substitution is contemplated in the foregoing disclosure and, in some instances, some features of the present invention may be employed without a corresponding use of the other features. Accordingly, it is appropriate that the foregoing description be construed broadly and understood as being given by way of illustration and example only, the spirit and scope of the invention being limited only by the appended claims.

**WHAT IS CLAIMED:**

1. A carrier-recovery loop that generates a reference signal with a reference frequency substantially equal to a received frequency of a received signal, the carrier-recovery loop comprising:
  - 5 a signal input that receives the received signal;
  - a digitally controlled oscillator (DCO) with an input and an output, wherein the output provides the reference signal that oscillates at the reference frequency;
  - a phase detector coupled to said signal input and to said DCO, wherein said phase detector receives the received signal from said signal input, wherein said phase detector receives the reference
  - 10 signal from said DCO, and wherein the phase detector generates a digital phase error signal, wherein the digital phase error signal indicates a phase difference between the received signal and the reference signal;
  - a memory that stores one or more gain coefficients; and
  - a loop filter coupled to said phase detector, to said memory, and to the input of said DCO, wherein:  - 15 said loop filter receives the digital phase error signal from said phase detector;
  - said loop filter receives the gain coefficients from said memory;
  - said loop filter operates on the digital phase error signal in response to the gain coefficients to generate a digital feedback signal;
  - wherein the digital feedback signal is provided to the input of said DCO, and wherein said
  - 20 DCO changes the reference frequency in response to the digital feedback signal.
2. The carrier-recovery loop of Claim 1, wherein the gain coefficients are programmable to place the carrier-recovery loop in one of a plurality of operating modes.
3. The carrier-recovery loop of Claim 2, wherein the plurality of operating modes includes a tracking mode, wherein in the tracking mode the gain coefficients have pre-programmed tracking values to allow the reference
- 25 signal to track the received signal.

4. The carrier-recovery loop of Claim 2, wherein the plurality of operating modes includes an acquisition mode, wherein in the acquisition mode the gain coefficients have pre-programmed acquisition values to enable the carrier recovery loop to acquire a phase lock with the received signal.

5. The carrier-recovery loop of Claim 2, wherein the plurality of operating modes includes a hold mode, wherein in the hold mode the gain coefficients are zero to prevent the reference signal from tracking the input signal.

6. The carrier-recovery loop of Claim 2, wherein the plurality of operating modes includes an acquisition mode and a hold mode, wherein:  
in the acquisition mode the gain coefficients have pre-programmed acquisition values to enable the carrier recovery loop to acquire a phase lock with the received signal; and

in the hold mode the gain coefficients are zero to prevent the reference signal from tracking the input signal.

7. The carrier-recovery loop of Claim 2, wherein the plurality of operating modes includes a tracking mode, an acquisition mode, and a hold mode, wherein:

15 in the tracking mode the gain coefficients have pre-programmed tracking values to allow the reference signal to track the received signal;

in the acquisition mode the gain coefficients have pre-programmed acquisition values to enable the carrier recovery loop to acquire a phase lock with the received signal; and

20 in the hold mode the gain coefficients are zero to prevent the reference signal from tracking the input signal.

8. The carrier-recovery loop of Claim 1, wherein said memory stores a first gain coefficient and a second gain coefficient, and wherein said loop filter further comprises:

25 a first multiplier coupled to said phase detector and to said memory that receives the digital phase error signal from said phase detector and the first gain coefficient from said memory, wherein said first multiplier multiplies the digital phase error signal by the first gain coefficient to generate a first product;

a second multiplier coupled to said phase detector and to said memory that receives the digital phase error signal from said phase detector and the second gain coefficient from said memory,

wherein said second multiplier multiplies the digital phase error signal by the second gain coefficient to generate a second product;

an integrator coupled to the second multiplier that receives the second product and integrates the second product to generate a digital integrated signal; and

5 an adder that adds the first product and the digital integrated signal to generate the digital feedback signal.

9. The carrier-recovery loop of Claim 1, wherein the digital phase error signal comprises a plurality of phase error bits, wherein the bits are a binary representation of the digital phase error signal, wherein the first and second gain coefficients are each powers of two; and wherein said first and second multipliers operate by  
10 shifting the digital phase error signal and digital integrated signal by an appropriate number of bits.

10. The carrier-recovery loop of Claim 1, wherein said DCO comprises:

a digital-to-analog converter coupled to said loop filter that receives the digital feedback signal and generates an analog feedback signal in response to the digital feedback signal; and

15 a voltage-controlled oscillator that receives the analog feedback signal from said digital-to-analog converter and generates the reference signal that oscillates at the reference frequency.

11. A method for performing carrier recovery in one of a plurality of selectable operating modes, the method comprising:

programming one or more gain coefficients of a loop filter to select one of the plurality of operating modes of the carrier recovery;

20 receiving a received signal, wherein the received signal has a received frequency;

locking a reference frequency of a reference signal to the received frequency of the received signal, wherein said locking comprises:

generating a digital phase error signal that represents a phase difference between the received signal and the reference signal;

25 filtering the digital phase error signal according to the programmed one or more gain coefficients to generate a digital feedback signal; and

adjusting the reference frequency of the reference signal in response to the digital feedback signal so that the reference frequency matches the received frequency.

12. The method of Claim 11, wherein the plurality of operating modes includes a tracking mode wherein in the tracking mode the gain coefficients have pre-programmed tracking values to allow the reference signal to track the received signal.

13. The method of Claim 11, wherein the plurality of operating modes includes an acquisition mode wherein in the acquisition mode the gain coefficients have pre-programmed acquisition values to enable the carrier recovery loop to acquire a phase lock with the received signal.

14. The method of Claim 11, wherein the plurality of operating modes includes a hold mode wherein in the hold mode the gain coefficients are zero to prevent the reference signal from tracking the input signal.

15. The method of Claim 11, wherein the plurality of operating modes includes an acquisition mode and a hold mode wherein:

in the acquisition mode the gain coefficients have pre-programmed acquisition values to enable the carrier recovery loop to acquire a phase lock with the received signal; and

in the hold mode the gain coefficients are zero to prevent the reference signal from tracking the input signal.

16. The method of Claim 11, wherein the plurality of operating modes includes a tracking mode, an acquisition mode, and a hold mode wherein:

in the tracking mode the gain coefficients have pre-programmed tracking values to allow the reference signal to track the received signal;

in the acquisition mode the gain coefficients have pre-programmed acquisition values to enable the carrier recovery loop to acquire a phase lock with the received signal;

in the hold mode the gain coefficients are zero to prevent the reference signal from tracking the input signal.

17. The method of Claim 11, wherein the one or more gain coefficients include a first gain coefficient and a second gain coefficient, wherein said filtering comprises:



- a first multiplying of the digital phase error signal by the first gain coefficient to generate a first product;
- a second multiplying of the digital phase error signal by the second gain coefficient to generate a second product;
- 5 integrating the second product in time to generate a digital integrated signal; and
- adding the first product and the digital integrated signal to generate the digital feedback signal.
18. The method of Claim 17, wherein the digital phase error signal comprises a plurality of phase error bits, wherein the bits are a binary representation of the digital phase error signal, wherein the first and second gain coefficients are each powers of two, wherein said first multiplying and said second multiplying comprise
- 10 shifting the digital phase error signal by an appropriate number of bits.
19. A loop filter adapted for operation in a carrier-recovery loop, wherein the carrier-recovery loop comprises a digitally controlled oscillator (DCO) and a phase detector, the loop filter comprising:
- a first multiplier coupled to the phase detector that receives a digital phase error signal from the phase detector, wherein said first multiplier multiplies the digital phase error signal by a first gain
- 15 coefficient to generate a first product;
- a second multiplier coupled to the phase detector that receives a digital phase error signal from the phase detector, wherein said second multiplier multiplies the digital phase error signal by a second gain coefficient to generate a second product;
- an integrator coupled to said second multiplier that receives the second product and integrates the
- 20 second product to generate a digital integrated signal; and
- an adder coupled to said first multiplier and to said integrator that adds the first product and the digital integrated signal to generate a digital feedback signal;
- wherein the digital feedback signal is provided to the input of the DCO, wherein the reference frequency of the DCO changes in response to the digital feedback signal.
20. The loop filter of Claim 19, wherein the first and second gain coefficients are programmable to place the carrier-recovery loop in one of a plurality of operating modes.

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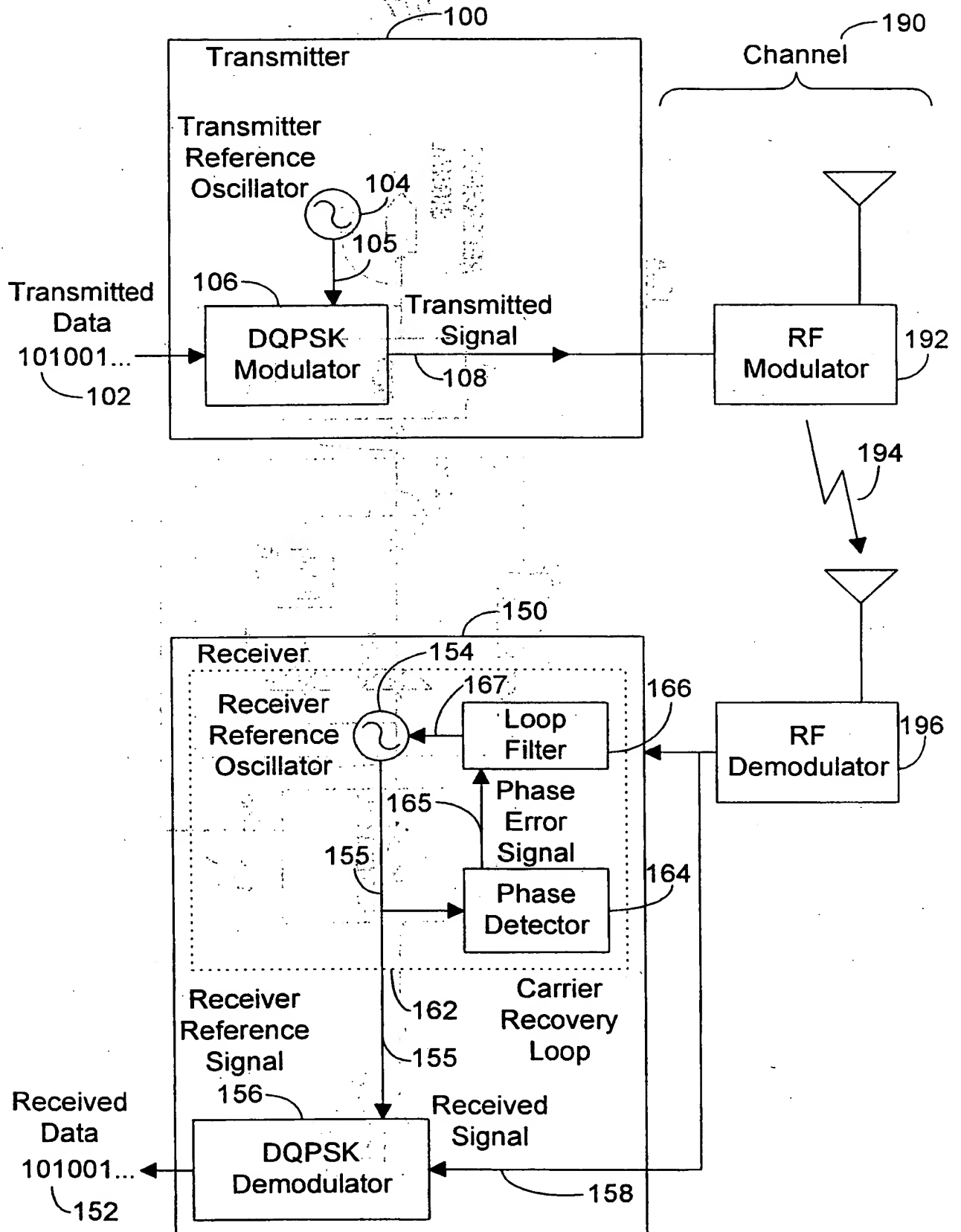


FIG. 1

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162A

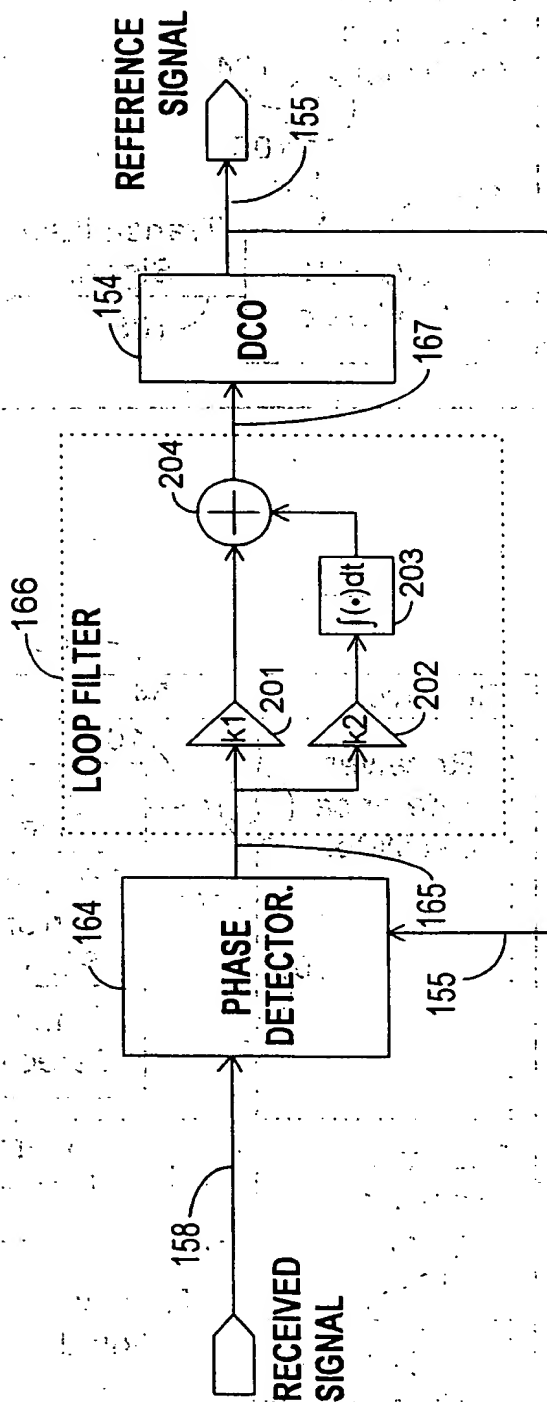


FIG. 2

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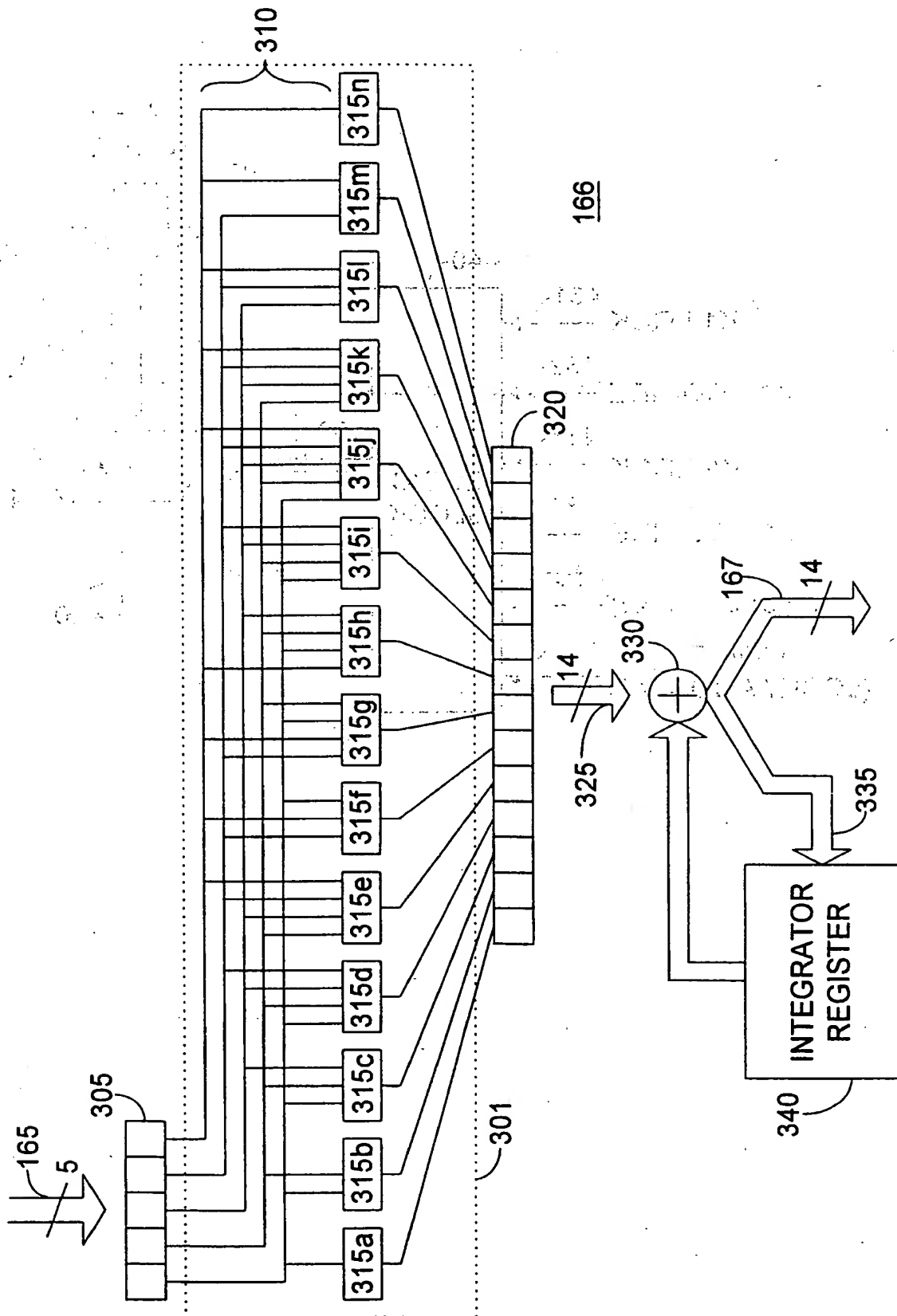


FIG. 3

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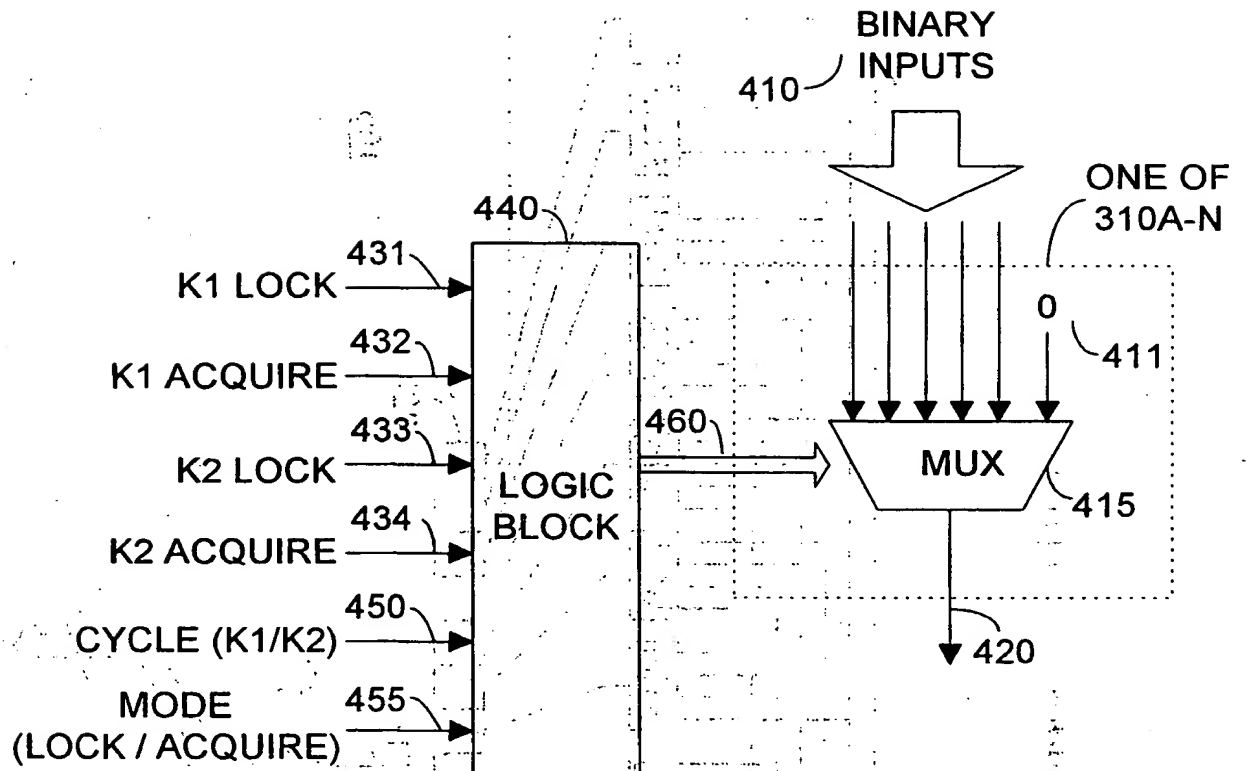


FIG. 4

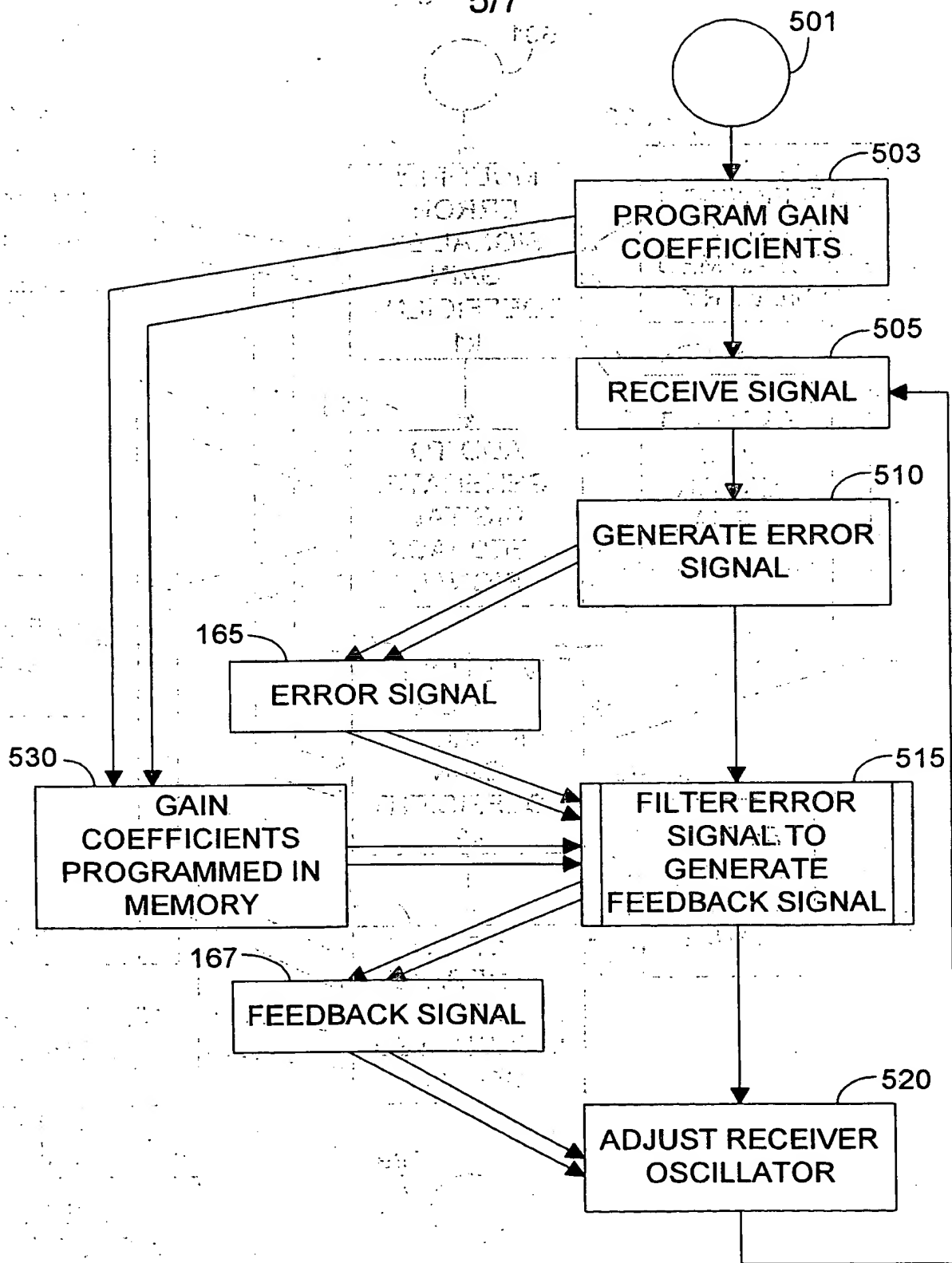


FIG. 5

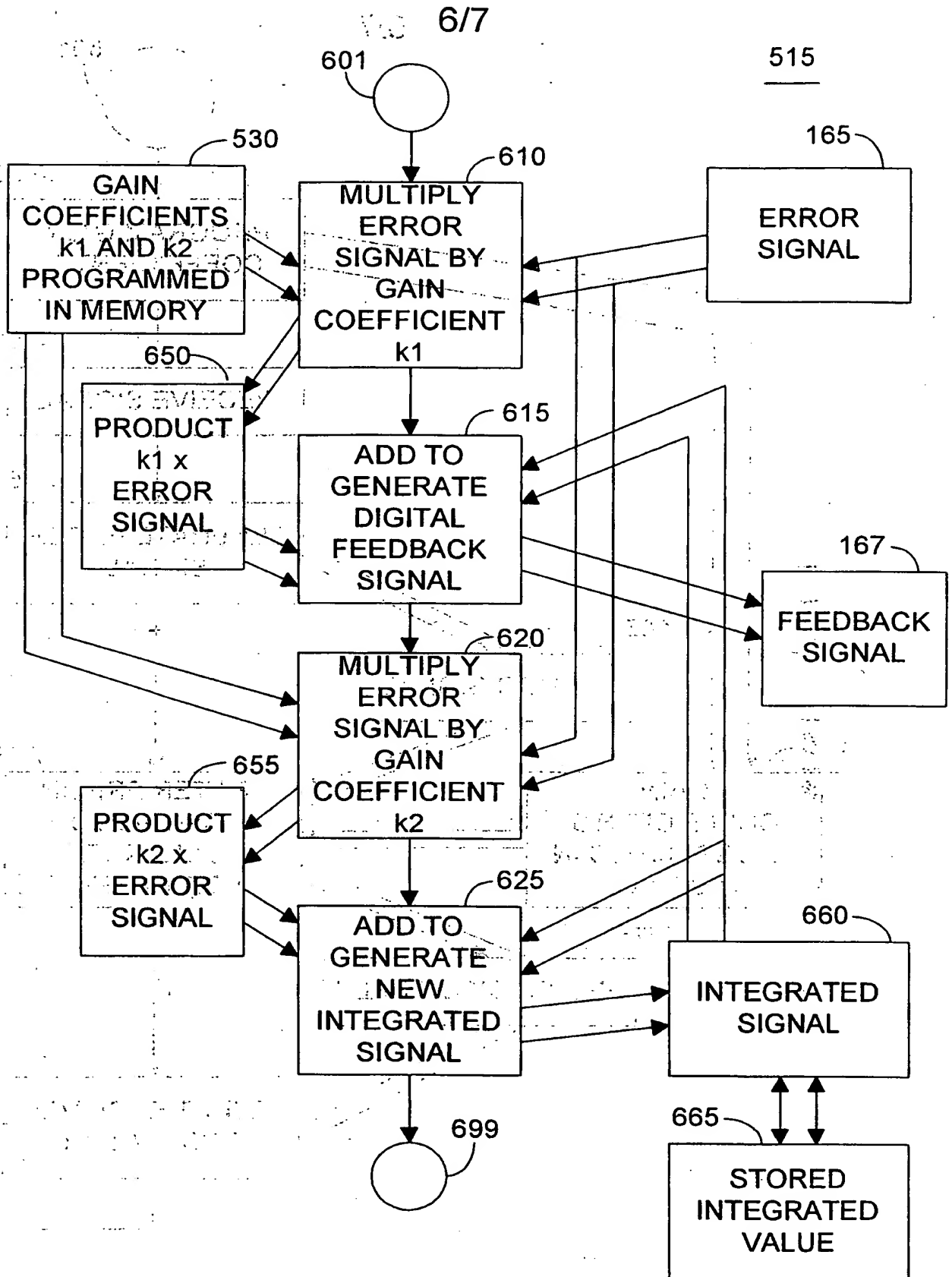


FIG. 6

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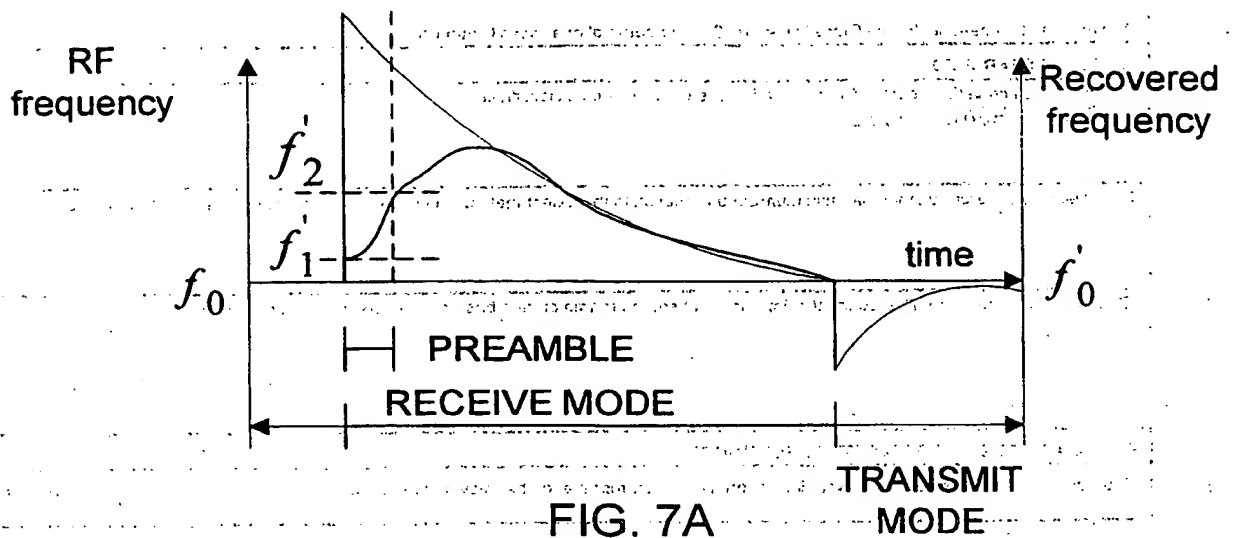


FIG. 7A

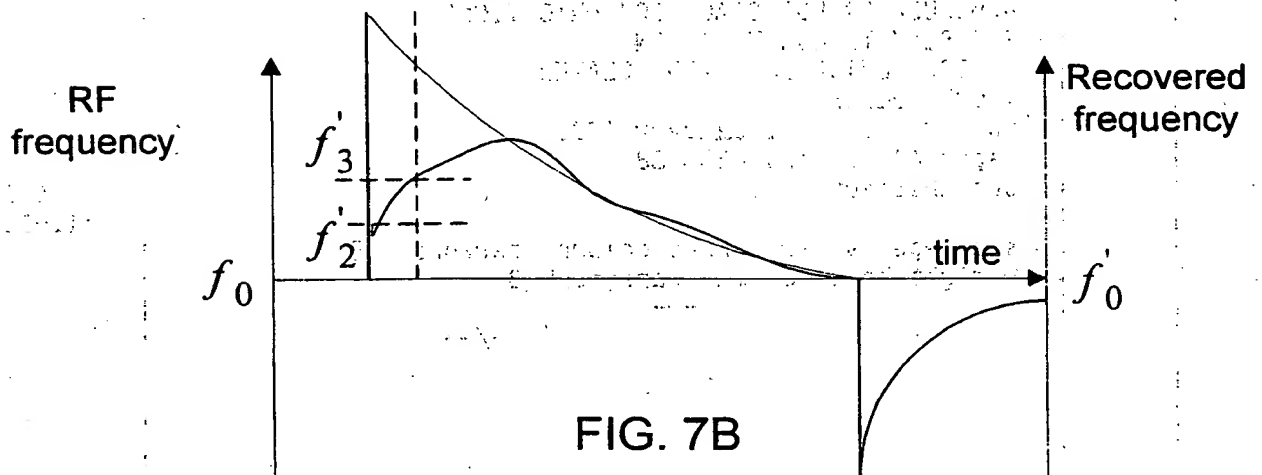


FIG. 7B

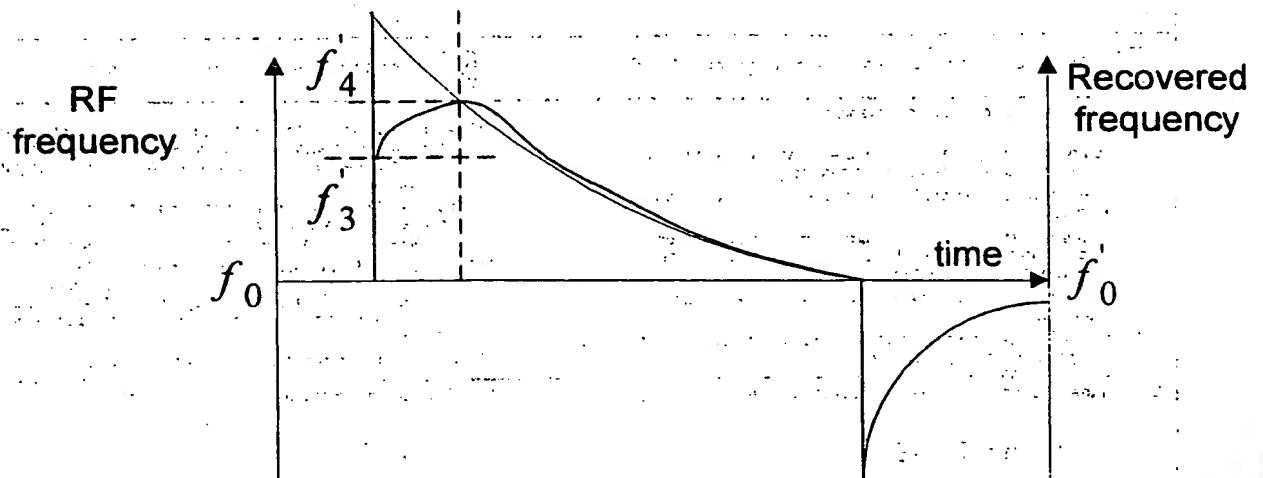


FIG. 7C



# INTERNATIONAL SEARCH REPORT

In International Application No

PCT/US 97/21469

**A. CLASSIFICATION OF SUBJECT MATTER**  
IPC 6 H04L27/227

According to International Patent Classification (IPC) or to both national classification and IPC

## B. FIELDS SEARCHED

Minimum documentation searched (classification system followed by classification symbols)

IPC: 6 H04L H03L

Documentation searched other than minimum documentation to the extent that such documents are included in the fields searched

Electronic data base consulted during the international search (name of data base and, where practical, search terms used)

## C. DOCUMENTS CONSIDERED TO BE RELEVANT

Category *	Citation of document, with indication, where appropriate, of the relevant passages	Relevant to claim No.
X	HANSEN F ET AL: "VLSI DIGITAL PSK DEMODULATOR FOR SPACE COMMUNICATION" EUROPEAN TRANSACTIONS ON TELECOMMUNICATIONS AND RELATED TECHNOLOGIES, vol. 4, no. 1, 1 January 1993, pages 43-52, XP000358888	19
Y	see abstract  see page 49, left-hand column, paragraph 5 - right-hand column, paragraph 3  -/-	1-4, 8, 11-13, 17

☒ Further documents are listed in the continuation of box C.

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Date of the actual completion of the international search

14 April 1998

Date of mailing of the international search report

27/04/1998

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## C. (Continuation) DOCUMENTS CONSIDERED TO BE RELEVANT

Category *	Citation of document, with indication, where appropriate, of the relevant passages	Relevant to claim No.
Y	UCHISHIMA M ET AL: "BURST DSP DEMODULATOR FOR LOW EB/NO OPERATION" COMMUNICATIONS - RISING TO THE HEIGHTS, DENVER, JUNE 23 - 26, 1991, vol. VOL. 1, no. -, 23 June 1991, INSTITUTE OF ELECTRICAL AND ELECTRONICS ENGINEERS, pages 226-230, XP000269407 see the whole document	1-4, 8, 11-13, 17
A	WO 96 12361 A (WESTINGHOUSE ELECTRIC CORP) 25 April 1996 see page 10, line 24 - line 26 see page 11, line 9 - line 28 see page 14, line 31 - page 16, line 17; figures 2,3	5-7
A	EP 0 590 323 A (SIEMENS AG ALBIS ;SIEMENS AG (DE)) 6 April 1994 see column 8, line 28 - column 9, line 9 see column 9, line 44 - column 10, line 6; figures 3-5	8,9
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Information on patent family members

International Application No

PCT/US 97/21469

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